

# TRANSMITTAL OF UTILITY PATENT APPLICATION FOR FILING

Certification under 37 CFR 1.10 (if applicable)

EL130481096US

January 27, 1999

"Express Mail" mailing label number

Date of Deposit

I hereby certify that this Transmittal letter, enclosed application, and any other documents referred to as enclosed herein, are being deposited in an envelope with the United States Postal Service "Express Mail Post Office to Addressee" service under 37 CFR 1.10 on the date indicated above and addressed to the Assistant Commissioner for Patents, Washington, D.C. 20231.

Cederic Rodgers

(Typed or printed name of person mailing application)

*Cederic Rodgers*

(Signature of person mailing application)

Box PATENT APPLICATION  
ASSISTANT COMMISSIONER FOR PATENTS  
Washington, D.C. 20231

Attorney Docket No. MM-99111

Customer Number 002387

Sir:

Transmitted herewith for filing is the utility patent application of inventor(s): **Ikuo Sunaga and Hiroyuki Matsumoto**

and entitled: **Transfer Scheme For Speech and Voice Band Signals and ISDN Digital Signals Using Reduced Transmission Bandwidth Over ATM**

1. Enclosed are:

- ☒ A duplicate copy of this transmittal letter.
- ☒ One stamped, self-addressed postcard for the PTO Mail Room date stamp.
- ☒ One utility patent application containing pages 1- 26, and
  - ☒ an executed declaration or oath for the utility patent application including a power of attorney,
  - ☐ an unexecuted declaration or oath for the utility patent application including a power of attorney, and
  - ☒ drawings:
    - ☒ 1 copy of 4 sheet(s) of formal drawings, OR
    - ☐ 1 copy of \_\_\_\_\_ sheet(s) of informal drawings, OR
    - ☐ 1 set of \_\_\_\_\_ Bristol board sheet(s) of original, formal drawings.
- ☒ A certified copy of Japanese application, No. P10-015968 filed January 28, 1998.
- ☐ An associate Power of Attorney.
- ☐ An Information Disclosure Statement.
- ☐ Verified Statement(s) relating to small entity status.
- ☐ Other: \_\_\_\_\_

2. The filing fee has been calculated as shown below:

	(Col. 1)	(Col. 2)
For:	Number Filed	Number Extra
Basic Fee		
Total Claims	14 — 20	= 0
Independent Claims	3 — 3	= 0
Multiple Dependent Claim Presented (if applicable)		

\* If the difference in Col. 1 is less than zero, enter "0" in Col. 2.

SMALL ENTITY	
Rate	Fee
	\$
x \$ 9 =	\$ -
x \$ 39 =	\$ -
+ \$130 =	\$ -
TOTAL	\$

OTHER THAN A SMALL ENTITY

Rate	Fee
	\$760.00
x \$ 18 =	\$--
x \$ 78 =	\$--
+ \$260 =	\$--
TOTAL	\$760.00

Please charge my Deposit Account No. 15-0508 in the amount of \$\_\_\_\_\_.

☒ A check in the amount of \$ 760.00 to cover the filing fee is enclosed.

TRANSMITTAL OF UTILITY PATENT APPLICATION FOR FILING

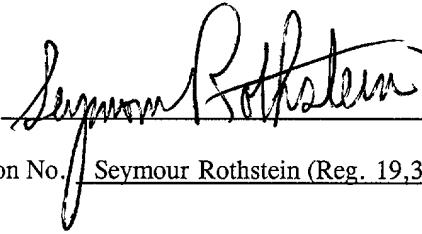
Page 2

- ☒ The Commissioner is authorized to charge payment of the following amounts associated with this communication or credit any overpayment to Deposit Account No. 15-0508:
- ☒ Additional filing fees under 37 CFR 1.16 or deficiencies in remittances therefor.
- ☒ Additional processing fees under 37 CFR 1.17 or deficiencies in remittances therefor.
- ☒ **ONLY if applicant has partially paid** the patent issue fee under 37 C.F.R. §1.18, then the **deficiency** shall be charged to Deposit Account No. 15-0508, and the Commissioner is authorized to so charge the Deposit Account.

3. The enclosed utility patent application is related to: Japanese Patent Application Serial No. P10-015968 filed January 28, 1998.

Date: January 27, 1999

Attorney's Signature



Name and Registration No. Seymour Rothstein (Reg. 19,369)

Correspondence Address:

OLSON & HIERL, LTD.  
20 North Wacker Drive  
36th Floor  
Chicago, Illinois 60606  
Telephone: (312) 580-1180

662270-01399

TRANSFER SCHEME FOR SPEECH AND VOICE BAND SIGNALS AND  
ISDN DIGITAL SIGNALS USING REDUCED TRANSMISSION  
BANDWIDTH OVER ATM

5

BACKGROUND OF THE INVENTION

FIELD OF THE INVENTION

The present invention relates to a transfer scheme for  
10 speech and voice band signals and ISDN (Integrated Services  
Digital Network) digital signals which transfers speech and  
voice band signals and ISDN digital signals entered from an  
STM (Synchronous Transfer Mode) network to an ATM  
(Asynchronous Transfer Mode) network by assembling ATM  
15 cells from them and transferring ATM cells entered from the  
ATM network to the STM network by converting them into STM  
signals, at a connection point between the ATM network and  
the STM network.

20 DESCRIPTION OF THE BACKGROUND ART

In the conventional STM network, signals are  
constantly transmitted through transmission channels  
regardless of presence or absence of communications such  
that unnecessary signals will occupy the transmission  
25 channels, and therefore it is impossible to realize an  
efficient utilization of transmission bandwidth. Also, at a  
time of transferring valid signals, the speech and voice  
band signals constantly occupy a certain transmission  
bandwidth even with respect to those signals which do not  
30 necessarily require the transmission bandwidth of one  
channel, so that it has been impossible to realize an  
efficient utilization of transmission channels.

Moreover, in the STM network in which signal  
compression is to be used, a compression scheme to be used  
35 is fixed, or the multiplexing of a plurality of signals

after the compression is carried out in units of an input channel group that are fixed in advance. In a DCME (Digital Circuit Multiplication Equipment) used as a speech compression equipment, it is impossible to judge a type of each received signal when compressed signals are received, so that there is a need to constantly carry out inter-equipment communications between transmitting and receiving DCMEs in order to make a mutual synchronization.

As described, conventionally, signals after the compression can be multiplexed only in units of an channel group that are fixed in advance, the transmitting network is always the same network, and the network quality for signals is always the same regardless of signal types so that it has been impossible to carry out separate communications depending on a signal type and it has been impossible to transfer signals of each signal type to a network suitable for its required quality.

Also, in the case of changing the compression scheme dynamically with respect to one communication call at the transmitting side using a plurality of compression schemes, there is a need to constantly carry out communications for control purpose between the transmitting side equipment and the receiving side equipment and make a synchronization between these equipments in order for the receiving side to recognize the compression scheme applied to each received signal, and it is necessary to provide a transmission bandwidth for this purpose.

## SUMMARY OF THE INVENTION

It is therefore an object of the present invention to provide a transfer scheme for speech and voice band signals and ISDN digital signals which is capable of reducing a transmission bandwidth of the ATM network required for

communications of speech, FAX, voice band data and ISDN digital signals that are to be entered from the STM network to the ATM network, and reducing the ATM cell assembling delay time.

5       According to one aspect of the present invention there is provided a transfer method for transferring speech and voice band signals and ISDN (Integrated Services Digital Network) digital signals between an ATM network and an STM (Synchronous Transfer Mode) network, the transfer method

10 comprising the steps of: (a) obtaining a silence information by detecting silence sections in input signals entered from the STM network; (b) obtaining a signal type information for each input signal by judging whether each input signal is a speech and voice band signal or an ISDN

15 digital signal and judging a signal type of the speech and voice band signal when each input signal is the speech and voice band signal; (c) dynamically changing a compression scheme of each input signal into a most appropriate compression scheme according to the silence information

20 obtained at the step (a) and the signal type information obtained at the step (b), and compressing each input signal using the most appropriate compression scheme; (d) assembling variable length packets each having a length shorter than that of an ATM cell from signals compressed at

25 the step (c) using the silence information obtained at the step (a) and the signal type information obtained at the step (b); (e) assembling ATM cells by multiplexing a plurality of the variable length packets assembled at the step (d), and transferring the ATM cells to the ATM

30 network; (f) receiving input ATM cells from the ATM network and disassembling the input ATM cells into received packets; (g) disassembling the received packets obtained at the step (f) into received signals; (h) judging a signal compression scheme of each received signal obtained at the

35 step (g); (i) expanding each received signal using the

signal compression scheme judged at the step (h); and (j) reproducing silence sections in signals expanded at the step (i) so as to generate STM signals, and transferring the STM signals to the STM network.

5       According to another aspect of the present invention there is provided a transmitting side device for transferring speech and voice band signals and ISDN (Integrated Services Digital Network) digital signals from an STM (Synchronous Transfer Mode) network to an ATM  
10 network, the device comprising: a silence detection unit for obtaining a silence information by detecting silence sections in input signals entered from the STM network; a signal type judgement unit for obtaining a signal type information for each input signal by judging whether each  
15 input signal is a speech and voice band signal or an ISDN digital signal and judging a signal type of the speech and voice band signal when each input signal is the speech and voice band signal; a signal compression unit for dynamically changing a compression scheme of each input  
20 signal into a most appropriate compression scheme according to the silence information obtained by the silence detection unit and the signal type information obtained by the signal type judgement unit, and compressing each input signal using the most appropriate compression scheme; a  
25 packet assembling unit for assembling variable length packets each having a length shorter than that of an ATM cell from signals compressed by the signal compression unit using the silence information obtained by the silence detection unit and the signal type information obtained by  
30 the signal type judgement unit; and an ATM cell assembling unit for assembling ATM cells by multiplexing a plurality of the variable length packets assembled by the packet assembling unit, and transferring the ATM cells to the ATM network.

35       According to another aspect of the present invention

there is provided a receiving side device for transferring speech and voice band signals and ISDN (Integrated Services Digital Network) digital signals from an ATM network to an STM (Synchronous Transfer Mode) network, the device

5 comprising: an ATM cell disassembling unit for receiving input ATM cells from the ATM network and disassembling the input ATM cells into received packets; a packet disassembling unit for disassembling the received packets obtained by the ATM cell disassembling unit into received  
10 signals; a signal compression scheme judgement unit for judging a signal compression scheme of each received signal obtained by the packet disassembling unit; a signal expansion unit for expanding each received signal using the signal compression scheme judged by the signal compression  
15 scheme judgement unit; and a silence reproduction unit for reproducing silence sections in signals expanded by the signal expansion unit so as to generate STM signals, and transferring the STM signals to the STM network.

Other features and advantages of the present invention  
20 will become apparent from the following description taken in conjunction with the accompanying drawings.

#### BRIEF DESCRIPTION OF THE DRAWINGS

25

Fig. 1 is a block diagram of a transmitting side device for realizing the transfer scheme over ATM for speech and voice band signals and ISDN digital signals according to one embodiment of the present invention.

30 Fig. 2 is a block diagram of a receiving side device for realizing the transfer scheme over ATM for speech and voice band signals and ISDN digital signals according to one embodiment of the present invention.

Fig. 3 is a flow chart for the operation of the  
35 transmitting side device of Fig. 1.

Fig. 4 is a flow chart for the operation of the receiving side device of Fig. 2.

## 5 DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENTS

Referring now to Fig. 1 to Fig. 4, one embodiment of the transfer scheme over ATM for speech and voice band signals and ISDN digital signals according to the present invention will be described in detail.

Fig. 1 and Fig. 2 respectively show a transmitting side device and a receiving side device for realizing the transfer scheme over ATM for speech and voice band signals and ISDN digital signals according to one embodiment of the present invention. More specifically, the transmitting side device of Fig. 1 assembles ATM cells from various signals such as speech and voice band signals including speech, FAX, and voice band data and ISDN digital signals which are entered from an STM network, and transfers these ATM cells to an ATM network, while the receiving side device of Fig. 2 receives the ATM cells from the transmitting side device through the ATM network, converts these ATM cells into STM signals such as the speech and voice band signals including speech, FAX, and voice band data and ISDN digital signals, and transfers these signals to the STM network.

As shown in Fig. 1, the transmitting side device generally comprises a signal processing function unit 1 and a data processing function unit 3.

The signal processing function unit 1 further comprises: a silence detection unit 11 for obtaining silence information detected in signals entered from the STM network; a signal type judgement unit 12 for judging whether each signal entered from the STM network is the speech and voice band signal or the ISDN digital signal, and further judging a signal type of the speech and voice



band signal in the case where entered signal is judged as the speech and voice band signal; and a signal compression unit 13 for dynamically changing a compression scheme of each signal from the STM network into a most appropriate compression scheme during communications according to the silence information detected by the silence detection unit 11 and the signal type information obtained by the signal type judgement unit 12, and compressing each entered signal using the most appropriate compression scheme.

Also, the data processing function unit 3 further comprises: a packet assembling unit 14 for assembling short packets such as AAL2 (ATM Adaptation Layer type 2) packets for example which are variable length packets shorter than the ATM cells, from the signals compressed by the signal compression unit 13, using the silence information from the silence detection unit 11 and the signal type information from the signal type judgement unit 12; and an ATM cell assembling unit 15 for assembling ATM cells from the short packets assembled by a plurality of packet assembling units 14, and outputting these ATM cells to the ATM network by multiplexing a plurality of ATM cells into a single ATM connection.

As shown in Fig. 2, the receiving side device generally comprises a data processing function unit 5 and a signal processing function unit 7.

The data processing function unit 5 further comprises: an ATM cell disassembling unit 21 for receiving ATM cells transferred from the transmitting side device of Fig. 1 to the ATM network and entered from the ATM network, and disassembling these ATM cells into packets; a packet disassembling unit 22 for disassembling packets generated by the ATM cell disassembling unit 21 so as to generate signals; and a signal compression scheme judgement unit 23 for judging a signal compression scheme of each signal generated by the packet disassembling unit 22.

Also, the signal processing function unit 7 further comprises: a signal expansion unit 24 for expanding signals using the signal compression scheme judged by the signal compression scheme judgement unit 23, and a silence reproduction unit 25 for reproducing silence sections in signals expanded by the signal expansion unit 24 so as to generate STM signals, and transferring these STM signals to the STM network.

The signal compression unit 13 of the transmitting side device includes a plurality of signal compression sub-units 13a, 13b, 13c, 13d and 13e, which respectively carry out signal compression using ADPCM (Adaptive Differential Pulse Code Modulation) scheme, LD-CELP (Low Delay-Code Excited Linear Prediction) scheme, CS-ACELP (Conjugate Structure-Algebraic Code Excited Linear Prediction) scheme, Modem demodulation scheme, and PCM (Pulse Code Modulation) scheme. The 64 kbit/s signals are compressed into 32 kbit/s signals by the ADPCM scheme, 16 kbit/s signals by the LD-CELP scheme, 8 kbit/s signals by the CS-ACELP scheme, while the Modem demodulation scheme is for FAX, where modulation is applied to the signals such that the signals can be demodulated at the receiving side because the FAX cannot be reproduced correctly at the receiving side if compressed, and the PCM scheme passes 64 kbit/s signals unchanged.

The packet assembling unit 14 of the transmitting side device includes a packet assembling sub-unit 14a for assembling packets from signals compressed by the ADPCM scheme, the LD-CELP scheme, the CS-ACELP scheme and the Modem demodulation scheme which are entered from the signal compression sub-units 13a, 13b, 13c and 13d, and a packet assembling sub-unit 14b for assembling packets from signals compressed by the PCM scheme which are entered from the signal compression sub-unit 13e.

The ATM cell assembling unit 15 of the transmitting side device forms ATM cells from the packets formed by a

plurality of packet assembling units 14, and transfers these ATM cells to the ATM network by multiplexing a plurality of ATM cells into a single ATM connection, where the multiplexing into a single ATM connection can be

5 carried out in units of an input channel group entered from the STM network or the multiplexing of the signals entered from the STM network into a single ATM connection can be carried out in units of a signal type judged by the signal type judgement unit 12 so that it is possible to freely

10 change a combination of data to be multiplexed such as the speech signals alone or the speech signals and FAX signals. Namely, the ATM cell assembling unit 15 has a plurality of multiplexing scheme such as the multiplexing in units of STM input channel group and the multiplexing in unit of

15 signal compression scheme and changes the multiplexing scheme among these plurality of multiplexing schemes at a time of transferring the ATM cells to the ATM network by multiplexing them into a single ATM connection.

On the other hand, the packet disassembling unit 22 of

20 the receiving side device has a packet disassembling sub-unit 22a and a packet disassembling sub-unit 22b in correspondence to the packet assembling sub-units 14a and 14b of the packet assembling unit 14, while the signal expansion units 24 of the receiving side device has signal

25 expansion sub-units 24a, 24b, 24c, 24d and 24e which respectively carry out signal expansion using ADPCM scheme, LD-CELP scheme, CS-ACELP scheme, Modem demodulation scheme, and PCM scheme, in correspondence to the signal compression sub-units 13a, 13b, 13c, 13d and 13e of the signal

30 compression unit 13.

The signal compression scheme judgement unit 23 of the receiving side device judges the signal compression type of each packet according to the packet length of the variable length packet, at a time of judging the signal type of the

35 short variable length packet within each ATM cell. Namely,

the packet assembling unit 14 of the transmitting side device uses packets with different packet lengths depending on data processed, so that it is possible to identify the signal compression scheme of each packet by detecting the packet length of each packet.

Next, with references to Fig. 3 and Fig. 4, the operations of the transmitting side device and the receiving side devices of Fig. 1 and Fig. 2 will be described.

First, with reference to Fig. 3, the operation of the transmitting side device will be described. One or a plurality of speech, FAX, and voice band data and ISDN digital signals entered from the STM network to the transmitting side device of Fig. 1 are supplied to the silence detection unit 11, the signal type judgement unit 12 and the signal compression unit 13. At the silence detection unit 11, the silence section in each received signal is detected, and the obtained silence information is notified to the packet assembling unit 14 and stored therein (step S11). Also, at the signal type judgement unit 12, a signal type of each received signal is judged, and the obtained signal type information is notified to the packet assembling unit 14 and stored therein (step S13). In this signal type judgement processing at the signal type judgement unit 12, whether the signal from the STM network is the speech and voice band signal or the ISDN digital signal is judged first, and when it is judged as the speech and voice band signal, a signal type of the speech and voice band signal is judged further.

Also, at the signal compression unit 13, the compression scheme of the input signal is dynamically changed during communications to the most appropriate compression scheme using the silence information obtained by the silence detection unit 11 and the signal type information obtained by the signal type judgement unit 12

such that the input signal is compressed by this most appropriate compression scheme, and the compressed data are entered and stored in the packet assembling unit 14 (step S15). The signal compression at the signal compression unit 5 13 is carried out by one of a plurality of signal compression sub-units 13a, 13b, 13c, 13d and 13e using the ADPCM scheme, LD-CELP scheme, the CS-ACELP scheme, the Modem demodulation scheme, and the PCM scheme, respectively.

10 The packet assembling unit 14 then carries out the signal compression scheme judgement processing (step S17) to judge the signal compression scheme of data stored at the step S15 according to the signal type information obtained from the signal type judgement unit 12. This 15 signal compression scheme is one of the ADPCM scheme, the LD-CELP scheme, the CS-ACELP scheme, the Modem demodulation scheme, and the PCM scheme described above.

In the ADPCM scheme data processing (step S19), the ending time of the silence section is detected by matching 20 the stored silence and speech information, the stored continuous data are segmented in units of a prescribed data length (11 octets, for example), and the packets are assembled. The segmentation using the same data length and the packet assembling are also applied to the subsequently 25 stored continuous data as well. Also, after attaching an information indicating the first non-silence section to a header of the short packet such as AAL2 packet, it is transmitted to the ATM cell assembling unit 15. Also, by matching the speech and silence information, those 30 assembled packets whose data section is entirely silence are discarded. In addition, for those assembled packets whose data section is non-silence, an immediately previous sequence counter value is incremented by one and attached to a header of the short packet, and these packets are 35 transmitted to the ATM cell assembling unit 15.

In the LD-CELP scheme data processing (step S21), the processing similar to the ADPCM scheme data processing of the step S19 is carried out except that the packet assembling is carried out using the packet length of the LD-CELP scheme (5 octets)  $\times$  n (where n is a positive integer).

In the CS-ACELP scheme data processing (step S23), the processing similar to the ADPCM scheme data processing of the step S19 is carried out except that the packet assembling is carried out using the packet length of the CS-ACELP scheme (10 octets)  $\times$  n (where n is a positive integer).

In the Modem demodulation scheme data processing (step S25) and the PCM scheme data processing (step S27), the processing similar to the ADPCM scheme data processing of the step S19 is carried out.

The packets generated in this way at the packet assembling unit 14, i.e., the short packets such as AAL2 packets, are then received and stored by the ATM cell assembling unit 15 (step S29). The ATM cell assembling unit 15 then applies the ATM multiplexing scheme judgement processing (step S31) to the stored short packets. In this ATM multiplexing scheme judgement processing, a scheme for mapping short packets to the ATM cell is judged for the stored short packets according to a prescribed one of a plurality of schemes for mapping to the ATM cell including the mapping per STM input channel group and the mapping per signal compression scheme (step S31). In the case where the judgement result is the mapping per STM input channel group, the STM input channel group judgement processing is carried out (step S33), and in the case where the judgement result is the mapping per signal compression scheme, the signal compression scheme judgement processing is carried out (step S41).

In the STM input channel group judgement processing in

the case of the mapping per STM input channel group, the corresponding ATM connection is judged for the short packets stored at the step S29 by referring to the ATM connection information set up for the corresponding channel group (step S35). Then, the stored short packets are mapped to the payload of the ATM cell on the corresponding one ATM connection by First In First Out policy for each input channel group of the STM network judged at the step S31 (step S37), and after the payload in the ATM cell is entirely filled by this mapping, the ATM cell is transmitted to the ATM network (step S39). Also, a timer is activated when the first short packet is mapped to each ATM cell and when this timer expires, even if the payload in the ATM cell is not entirely filled, an invalid data is given to the payload in the ATM cell where no data has been mapped, and the ATM cell is transmitted to the ATM network (step S39).

On the other hand, in the signal compression scheme judgement processing in the case of the mapping per signal compression scheme, the corresponding ATM connection is judged for the short packets stored at the step S29 by referring to the ATM connection information set up for the corresponding signal compression scheme (step S43). Then, the stored short packets are mapped to the payload of the ATM cell on the corresponding one ATM connection by First In First Out Policy for each signal compression scheme judged at the step S41 (step S45), and after the payload in the ATM cell is entirely filled by this mapping, the ATM cell is transmitted to the ATM network (step S47). Also, a timer is activated when the first short packet is mapped to each ATM cell and when this timer expires, even if the payload in the ATM cell is not entirely filled, an invalid data is given to the payload in the ATM cell where no data has been mapped, and the ATM cell is transmitted to the ATM network (step S47).

Next, with reference to Fig. 4, the operation of the receiving side device of Fig. 2 will be described. The ATM cell entered from the ATM network at the receiving side device of Fig. 2 is received by the ATM cell disassembling unit 21 and disassembled from 53 bytes ATM cell into short packets (step S51). Then, according to the short packet connection identifier field value in each disassembled short packet, each short packet is transferred to the packet disassembling unit 22 of a channel which is in one-to-one correspondence to a channel on the STM side which is specified in advance to that short packet connection identifier value (the short packet identifier 1 = STM side channel number 1, the short packet identifier value 2 = STM side channel number 2, and so on, for example) (step S53).

The packet disassembling unit 22 then receives and stores the short packet transferred from the ATM cell disassembling unit 21 (step S55). Then, the first packet of the continuous data, or the intermediate packet of the continuous data, is detected according to the UUI (User-User Information) field value in the short packet header of this short packet (step S57).

More specifically, this packet detection can be realized as follows. At the transmitting side for transferring the packet to the ATM network, the packet assembling unit 14 eliminates the silence section detected by the silence detection unit 11, assembles packets only from the non-silence section, and transmits the packets to the ATM cell assembling unit 15 while using the sequence counter ranging from 0 to 7 in the headers of the short packets in this transmission such that 0 is used for the first short packet of the non-silence section and 1 to 7 are repeatedly used for the subsequent consecutive short packets until the silence section occurs, and the sequence counter is reset when the silence section occurs such that 0 is used for the first short packet of the next non-



silence section and 1 to 7 are repeatedly used for the subsequent consecutive short packets.

Then, at the ATM cell receiving side, the short packets assembled only from the non-silence section by  
5 eliminating the silence section are received, while the sequence counter value in the headers of the short packets are detected in the processing for receiving the short packets in the ATM cell such that the short packet for which the sequence counter value is 0 is detected as the  
10 first packet of the non-silence section and the short packets for which the sequence counter values are 1 to 7 are detected as the packets of the consecutive non-silence section. In this way, the first short packet of the non-silence section is detected, and if there is any loss of  
15 the consecutive short packets of the non-silence section between the transmitting side and the receiving side, the occurrence of the loss as well as the number of lost short packets are detected.

Then, each silence section is detected as a no signal  
20 period preceding each packet with the sequence counter value 0, and the silence information indicating the detected silence sections is notified to the signal expansion unit 24 and the silence reproduction unit 25 (step S59). Then, the packet disassembling unit 22 removes  
25 the short packet headers and transmits the remaining valid data packets to the signal compression scheme judgement unit 23 (step S61).

The signal compression scheme judgement unit 23 receives and stores the valid data packets transferred from  
30 the packet disassembling unit 22 (step S63), and detects the packet length of the stored valid data packets (step S65). Then, the signal compression scheme judgement unit 23 judges the signal expansion scheme of the stored valid data packets by matching the packet length of the stored valid  
35 data packets with the packet length uniquely specified for

each of the ADPCM scheme, the LD-CELP scheme, the CS-ACELP scheme, the Modem demodulation scheme, and the PCM scheme available in the signal expansion unit 24 (n octets, where n is a positive integer: 11 octets for the ADPCM scheme, 5 octets for the LD-CELP scheme, 10 octets for the CS-ACELP scheme, 9 octets for the Modem demodulation scheme, and 20 octets for the PCM scheme, for example) (step S67).

The signal compression scheme judgement unit 23 then transmits the valid data packets to the signal expansion sub-units 24a to 24e corresponding to the signal expansion schemes so judged (step S69). The signal expansion sub-units 24a to 24e of the signal expansion unit 24 then carries out the expansion processing for the valid data using the ADPCM scheme, the LD-CELP scheme, the CS-ACELP scheme, the Modem demodulation scheme, and the PCM scheme respectively (step S71), and supply the expanded data to the silence reproduction unit 25.

Here, the signal expansion sub-units 24a to 24e of the signal expansion unit 24 apply the most appropriate loss compensation scheme when the loss of packet is encountered during the expansion processing in such a way that, when the received packet with the sequence counter value 0 is detected, the signal expansion unit 24 resets an algorithm of the signal compression scheme by taking the received packet with the sequence counter value 0 as the first packet of the non-silence section, so as to enable improvement of a quality of reproduced speech and voice band signals, and when the loss of packet is detected, the signal expansion unit 24 makes a judgement as to whether a lost packet is the first packet of the non-silence section or one of the subsequent packets of the non-silence section, and applies a most appropriate loss compensation scheme according to a result of the judgement.

Note that the loss of the first packet of the non-silence section can be identified by detecting an

occurrence of packets with the sequence counter values other than 0 which are immediately preceded not by a packet with the sequence counter value 0 but by a no signal period longer than a prescribed length.

5       Note also that the most appropriate loss compensation scheme for each case can be any known loss compensation scheme such as those described in the ITU-U recommendation G.728 and G.729, for example.

10       The silence reproduction unit 25 reproduces the silence sections in the expanded data from the signal expansion unit 24 according to the silence information notified from the packet disassembling unit 22 at the step S59, and transfers the resulting data as the STM signals to the STM network (step S73).

15

As described, according to the present invention, in the ATM cell multiplexing at the transmitting side, the signal type judgement results for the signals entered from the STM network are notified to the ATM cell multiplexing unit, and these signals are multiplexed into the ATM cell according to the notified signal types so that it becomes possible to multiplex the signals for each input signal type separately, and it becomes possible to treat the STM signals with different network qualities for different signal types as a single ATM connection in which only the input signals with the same quality requirement are multiplexed within the ATM network. Also, it becomes possible to reduce the transmission bandwidth required for communications as well as the ATM cell assembling delay time. In addition, by attaching the channel number identifiers corresponding to the STM network channels at the transmitting side, it becomes possible for the receiving side to identify corresponding channels on the ATM network side from the received ATM cells or variable length signal packets.

20  
25  
30  
35

Also, according to the present invention, the signal compression scheme applied at the transmitting side is judged according to the packet length of the variable length packet in the ATM cell received from the ATM network, so that it becomes possible to judge the signal compression scheme applied by the transmitting side equipment by setting the packet length of the variable length packet and the signal compression scheme in correspondence in advance. Consequently, an identifier for identifying the signal compression scheme in the ATM cell becomes unnecessary so that the utilization efficiency of the payload in the ATM cell can be improved.

In addition, according to the present invention, a plurality of schemes for multiplexing into the ATM connection including the multiplexing per STM input channel group and the multiplexing per signal compression scheme can be provided and the multiplexing scheme can be changed freely among these plurality of multiplexing schemes.

It is to be noted that, besides those already mentioned above, many modifications and variations of the above embodiment may be made without departing from the novel and advantageous features of the present invention. Accordingly, all such modifications and variations are intended to be included within the scope of the appended claims.

WHAT IS CLAIMED IS:

1. A transfer method for transferring speech and voice band signals and ISDN (Integrated Services Digital Network) digital signals between an ATM (Asynchronous Transfer Mode) network and an STM (Synchronous Transfer Mode) network, the transfer method comprising the steps of:
- (a) obtaining a silence information by detecting silence sections in input signals entered from the STM network;
  - 10 (b) obtaining a signal type information for each input signal by judging whether each input signal is a speech and voice band signal or an ISDN digital signal and judging a signal type of the speech and voice band signal when each input signal is the speech and voice band signal;
  - 15 (c) dynamically changing a compression scheme of each input signal into a most appropriate compression scheme according to the silence information obtained at the step (a) and the signal type information obtained at the step (b), and compressing each input signal using the most
  - 20 appropriate compression scheme;
  - (d) assembling variable length packets each having a length shorter than that of an ATM cell from signals compressed at the step (c) using the silence information obtained at the step (a) and the signal type information
  - 25 obtained at the step (b);
  - (e) assembling ATM cells by multiplexing a plurality of the variable length packets assembled at the step (d), and transferring the ATM cells to the ATM network;
  - (f) receiving input ATM cells from the ATM network and
  - 30 disassembling the input ATM cells into received packets;
  - (g) disassembling the received packets obtained at the step (f) into received signals;
  - (h) judging a signal compression scheme of each received signal obtained at the step (g);
  - 35 (i) expanding each received signal using the signal

compression scheme judged at the step (h); and

(j) reproducing silence sections in signals expanded at the step (i) so as to generate STM signals, and transferring the STM signals to the STM network.

5

2. The transfer method of claim 1, wherein the step (h) judges the signal compression scheme of each received signal obtained from each received packet according to a packet length of each received packet which is a variable  
10 length packet.

3. The transfer method of claim 1, wherein the step (e) assembles the ATM cells by multiplexing the variable length packets into an ATM connection in units of STM input  
15 channel group.

4. The transfer method of claim 1, wherein the step (e) assembles the ATM cells by multiplexing the variable length packets into an ATM connection in units of a signal type  
20 indicated by the signal type information obtained at the step (b).

5. The transfer method of claim 1, wherein the step (d) at an ATM cell transmitting side assembles the variable  
25 length packets only from non-silence sections by eliminating silence sections indicated by the silence information from signals compressed at the step (c), while using a sequence counter with a value ranging from 0 to 7 in headers of the variable length packets such that a value  
30 0 is used for a first packet of each non-silence section, values 1 to 7 are used repeatedly for subsequent packets of each non-silence section, and a sequence counter is reset when a silence section occurs;

the step (f) at an ATM cell receiving side receives  
35 the input ATM cells containing the received packets that

are assembled only from non-silence sections by eliminating  
silence sections, and the step (g) detects a sequence  
counter value in a header of each received packet such that  
a received packet with the sequence counter value 0 is  
5 detected as a first packet of each non-silence section and  
received packets with the sequence counter values 1 to 7  
are detected as subsequent packets of each non-silence  
section, so as to detect the first packet of each non-  
silence section as well as an occurrence of a loss of  
10 packets and a number of lost packets for each non-silence  
section when there is a loss of any consecutive packets of  
each non-silence section between a transmitting side and a  
receiving side; and

when the received packet with the sequence counter  
15 value 0 is detected, the step (i) resets an algorithm of  
the signal compression scheme by taking the received packet  
with the sequence counter value 0 as the first packet of  
the non-silence section, so as to enable improvement of a  
quality of reproduced speech and voice band signals, and  
20 when a loss of packet is detected, the step (i) makes a  
judgement as to whether a lost packet is the first packet  
of the non-silence section or one of the subsequent packets  
of the non-silence section, and applies a most appropriate  
loss compensation scheme according to a result of the  
25 judgement.

6. The transfer method of claim 1, wherein the step (e)  
changes a scheme for multiplexing a plurality of the  
variable length packets into an ATM connection among a  
30 plurality of schemes for multiplexing into an ATM  
connection including a multiplexing in units of STM input  
channel group and a multiplexing in units of signal  
compression scheme.

35 7. A transmitting side device for transferring speech and

voice band signals and ISDN (Integrated Services Digital Network) digital signals from an STM (Synchronous Transfer Mode) network to an ATM (Asynchronous Transfer Mode) network, the device comprising:

5        a silence detection unit for obtaining a silence information by detecting silence sections in input signals entered from the STM network;

10       a signal type judgement unit for obtaining a signal type information for each input signal by judging whether each input signal is a speech and voice band signal or an ISDN digital signal and judging a signal type of the speech and voice band signal when each input signal is the speech and voice band signal;

15       a signal compression unit for dynamically changing a compression scheme of each input signal into a most appropriate compression scheme according to the silence information obtained by the silence detection unit and the signal type information obtained by the signal type judgement unit, and compressing each input signal using the  
20       most appropriate compression scheme;

25       a packet assembling unit for assembling variable length packets each having a length shorter than that of an ATM cell from signals compressed by the signal compression unit using the silence information obtained by the silence detection unit and the signal type information obtained by the signal type judgement unit; and

30       an ATM cell assembling unit for assembling ATM cells by multiplexing a plurality of the variable length packets assembled by the packet assembling unit, and transferring the ATM cells to the ATM network.

8.       The device of claim 7, wherein the ATM cell assembling unit assembles the ATM cells by multiplexing the variable length packets in units of STM input channel group into an  
35       ATM connection.



9. The device of claim 7, wherein the ATM cell assembling unit assembles the ATM cells by multiplexing the variable length packets in units of a signal type indicated by the signal type information obtained by the signal type judgement unit into an ATM connection.

10. The device of claim 7, wherein the packet assembling unit assembles the variable length packets only from non-silence sections by eliminating silence sections indicated by the silence information from signals compressed by the signal compression unit, while using a sequence counter with a value ranging from 0 to 7 in headers of the variable length packets such that a value 0 is used for a first packet of each non-silence section, values 1 to 7 are used repeatedly for subsequent packets of each non-silence section, and a sequence counter is reset when a silence section occurs.

11. The device of claim 7, wherein the ATM cell assembling unit changes a scheme for multiplexing a plurality of the variable length packets into an ATM connection among a plurality of schemes for multiplexing into an ATM connection including a multiplexing in units of STM input channel group and a multiplexing in units of signal compression scheme.

12. A receiving side device for transferring speech and voice band signals and ISDN (Integrated Services Digital Network) digital signals from an ATM (Asynchronous Transfer Mode) network to an STM (Synchronous Transfer Mode) network, the device comprising:

an ATM cell disassembling unit for receiving input ATM cells from the ATM network and disassembling the input ATM cells into received packets;

a packet disassembling unit for disassembling the received packets obtained by the ATM cell disassembling unit into received signals;

5 a signal compression scheme judgement unit for judging a signal compression scheme of each received signal obtained by the packet disassembling unit;

a signal expansion unit for expanding each received signal using the signal compression scheme judged by the signal compression scheme judgement unit; and

10 a silence reproduction unit for reproducing silence sections in signals expanded by the signal expansion unit so as to generate STM signals, and transferring the STM signals to the STM network.

15 13. The device of claim 12, wherein the signal compression scheme judgement unit judges the signal compression scheme of each received signal obtained from each received packet according to a packet length of each received packet which is a variable length packet.

20 14. The device of claim 12, wherein the ATM cell disassembling unit receives the input ATM cells containing the received packets that are assembled only from non-silence sections by eliminating silence sections, and  
25 the packet disassembling unit detects a sequence counter value in a header of each received packet such that a received packet with the sequence counter value 0 is detected as a first packet of each non-silence section and received packets with the sequence counter values 1 to 7  
30 are detected as subsequent packets of each non-silence section, so as to detect the first packet of each non-silence section as well as an occurrence of a loss of packets and a number of lost packets for each non-silence section when there is a loss of any consecutive  
35 packets of each non-silence section between a transmitting

side and a receiving side; and

when the received packet with the sequence counter value 0 is detected, the signal expansion unit resets an algorithm of the signal compression scheme by taking the  
5 received packet with the sequence counter value 0 as the first packet of the non-silence section, so as to enable improvement of a quality of reproduced speech and voice band signals, and when a loss of packet is detected, the signal expansion unit makes a judgement as to whether a  
10 lost packet is the first packet of the non-silence section or one of the subsequent packets of the non-silence section, and applies a most appropriate loss compensation scheme according to a result of the judgement.

15

20

25

30

35

ABSTRACT OF THE DISCLOSURE

A transfer scheme for speech and voice band signals and ISDN digital signals which is capable of reducing a transmission bandwidth required for communications of speech, FAX, voice band data and ISDN digital signals that are to be entered from the STM network to the ATM network, and reducing the ATM cell assembling delay time, is disclosed. In the ATM cell multiplexing at the transmitting side, the signal type judgement results for the signals entered from the STM network are notified to the ATM cell multiplexing unit, and these signals are multiplexed into the ATM cell according to the notified signal types so that it becomes possible to multiplex the signals for each input signal type separately, and it becomes possible to treat the STM signals with different network qualities for different signal types as a single ATM connection in which only the input signals with the same quality requirement are multiplexed within the ATM network.

FIG.1

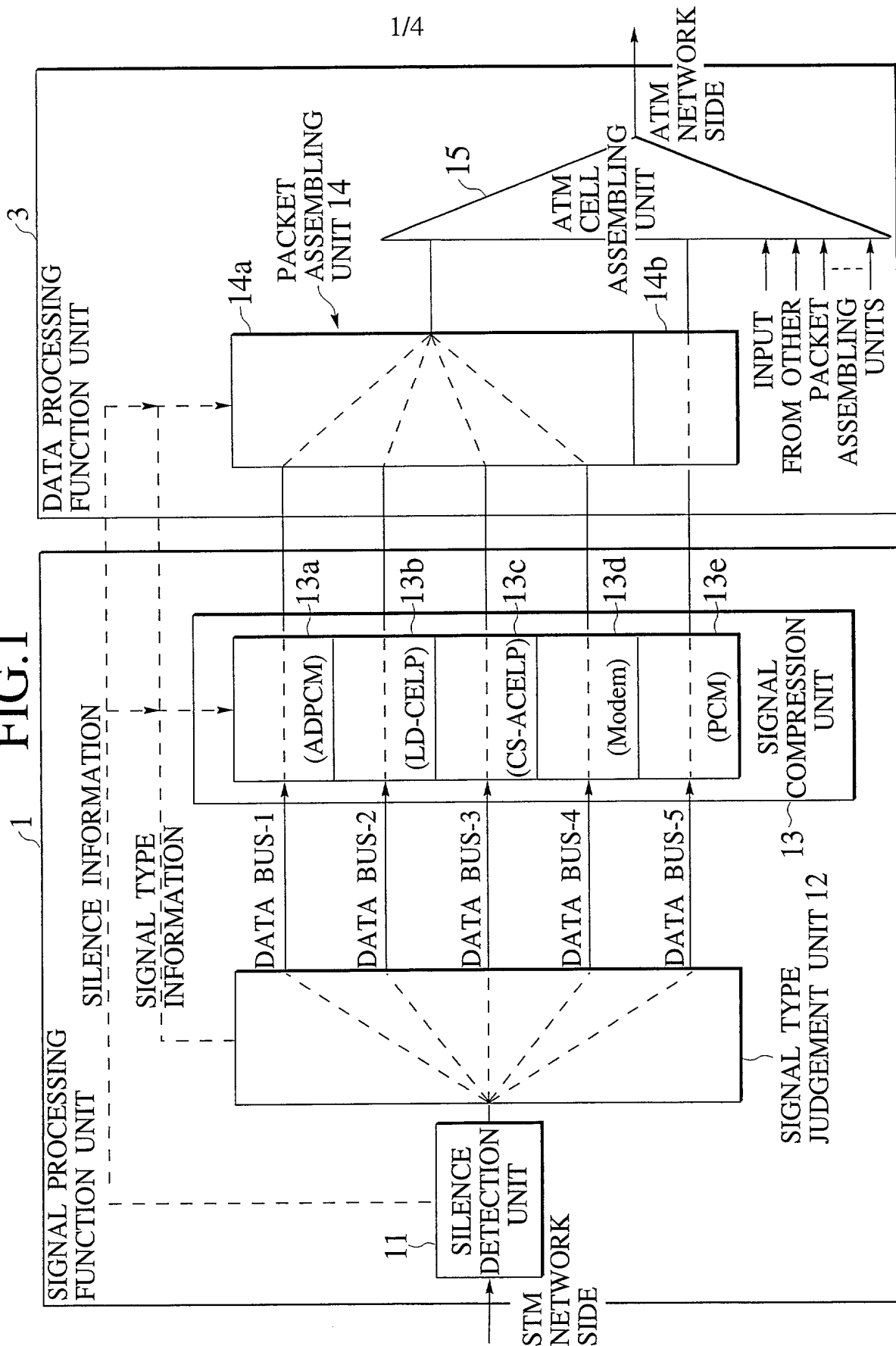


FIG. 2

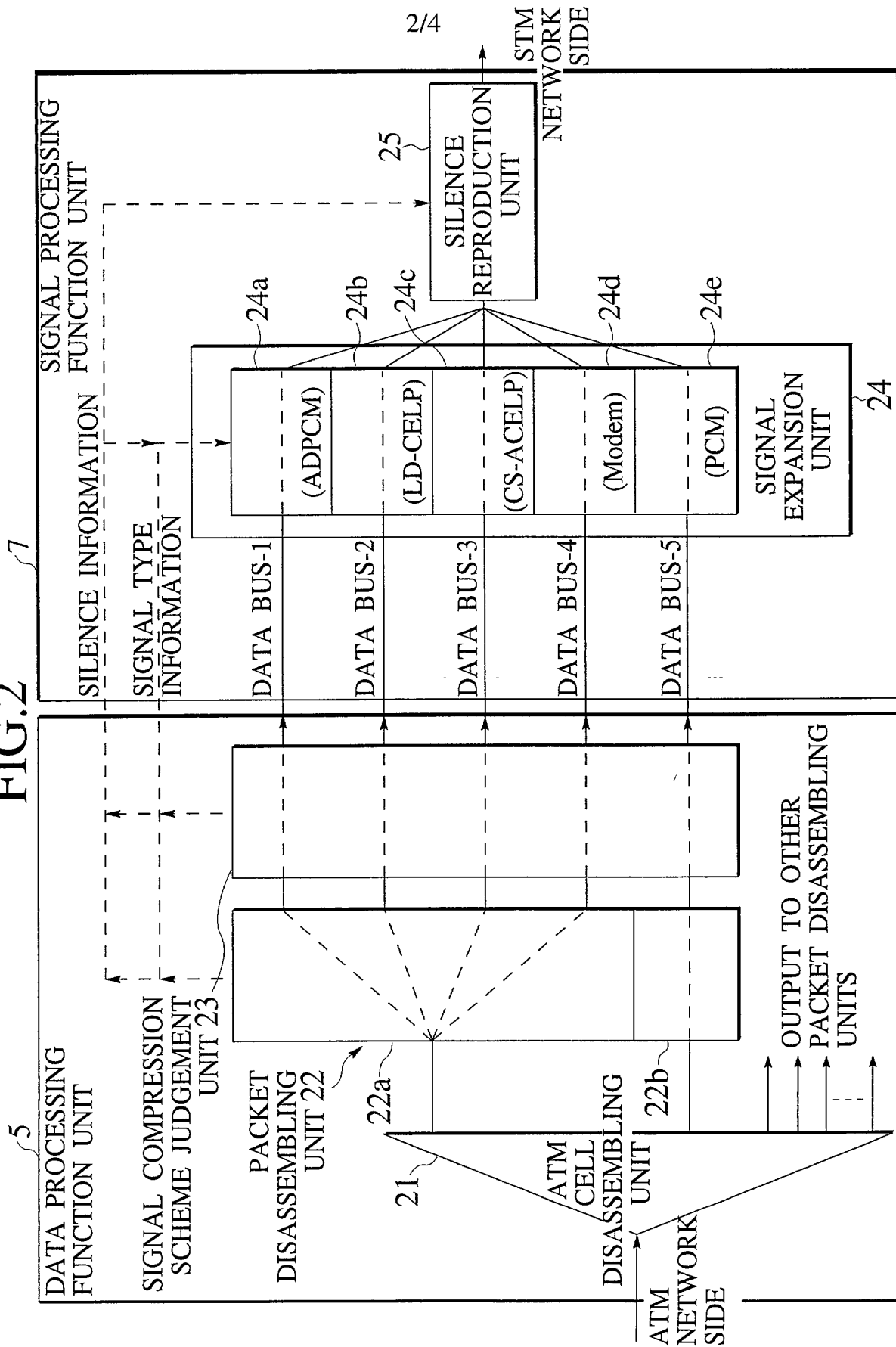


FIG.3

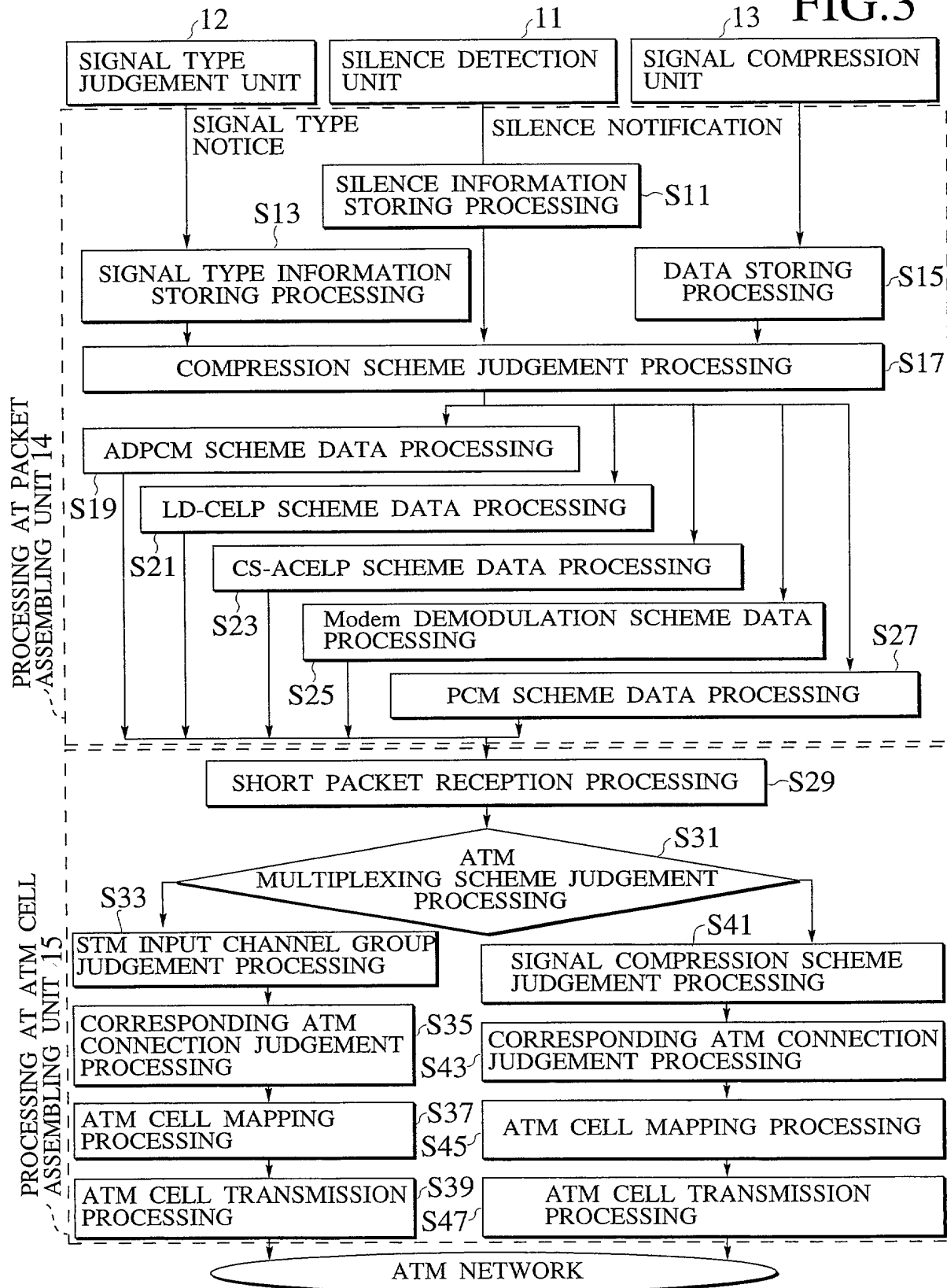
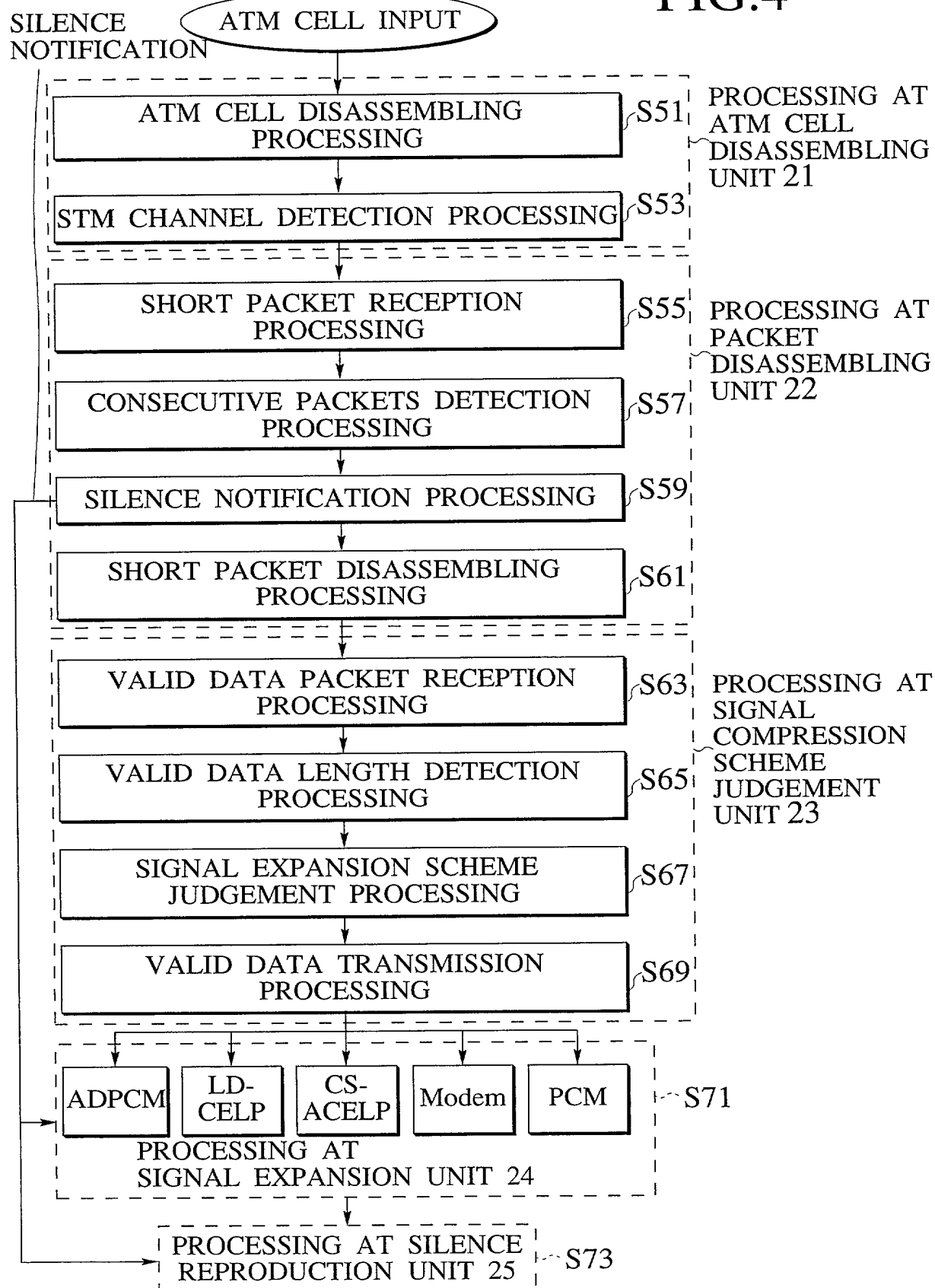


FIG.4





## PATENT APPLICATION DECLARATION AND POWER OF ATTORNEY

I HEREBY DECLARE THAT:

My residence, post office address, and citizenship are as stated next to my name in PART A on page 2 hereof.

I believe I am the original, first, and sole inventor (if only one name is listed) or an original, first, and joint inventor (if plural names are listed) of the subject matter which is claimed and for which a patent is sought on the invention entitled: TRANSFER SCHEME FOR SPEECH AND VOICE BAND SIGNALS AND ISDN DIGITAL SIGNALS USING REDUCED TRANSMISSION BANDWIDTH OVER ATM, the specification of which:

X is attached hereto;

\_\_\_\_\_ was filed on \_\_\_\_\_ as Application Serial No. \_\_\_\_\_ and was amended on \_\_\_\_\_ (if applicable).

I hereby state that I have reviewed and understand the contents of the above-identified specification, including the claims, as amended by any amendment referred to herein.

I acknowledge the duty to disclose information to the Patent and Trademark Office known to me to be material to the patentability of this application, as defined in Title 37, Code of Federal Regulations, Sec. 1.56.

I hereby claim foreign priority benefits under Title 35, United States Code, Sec. 119 of any foreign application(s) for patent or inventor's certificate listed in PART B on page 2 hereof and have also identified in PART B on page 2 hereof any foreign application for patent or inventor's certificate having a filing date before that of the application on which priority is claimed.

I hereby claim the benefit under Title 35, United States Code, Sec. 120 of any United States application(s) listed in PART C of page 2 hereof and, insofar as the subject matter of each of the claims of this application is not disclosed in the prior United States application in the manner provided by the first paragraph of Title 35, United States Code, Sec. 112, I acknowledge the duty to disclose all information to the Patent and Trademark Office known to me to be material to patentability of this application, as defined in Title 37, Code of Federal Regulations, Sec. 1.56, which became available between the filing date of the prior application and the national or PCT international filing date of this application.

I hereby declare that all statements made herein of my knowledge are true and that all statements made on information and belief are believed to be true; and further that these statements were made with the knowledge that willful false statements and the like so made are punishable by fine or imprisonment, or both, under Section 1001 of Title 18 of the United States Code and that such willful false statements may jeopardize the validity of the application or any patent issued thereon.

I hereby appoint the following as my attorneys or agents with full power of substitution to prosecute this application and transact all business in the United States Patent and Trademark Office connected therewith:

Michael A. Hierl	Reg. No. 29,807	Arne M. Olson	Reg. No. 30,203
Dolores T. Kenney	Reg. No. 31,269	Talivaldis Cepuritis	Reg. No. 20,818
John W. Klooster	Reg. No. 18,953	Daniel J. Deneufbourg	Reg. No. 33,675
Seymour Rothstein	Reg. No. 19,369	Eddie L. Bishop	Reg. No. 39,110

whose mailing address for this application is: OLSON & HIERL, LTD.  
20 North Wacker Drive  
36th Floor  
Chicago, Illinois 60606  
Telephone: (312) 580-1180

See Page 2 attached, signed, and made a part hereof.

9-1540

29

## PATENT APPLICATION DECLARATION AND POWER OF ATTORNEY

### PART A: Inventor Information and Signature

Full name of SOLE or FIRST inventor Ikuo SUNAGA  
Citizenship Japan Residence 1-14-12-504, Sakurajyosui, Setagaya-ku,  
Tokyo, Japan

Post Office Address (If different) \_\_\_\_\_

Inventor's signature: Ikuo Sunaga Date: January 22, 1999

Full name of SECOND joint inventor, if any Hiroyuki MATSUMOTO  
Citizenship Japan Residence 3-20-10, Tsukagoshi, Warabi-shi,  
Saitama-ken, Japan

Post Office Address (If different) \_\_\_\_\_

Second Inventor's signature: Hiroyuki Matsumoto Date: January 22, 1999

Full name of THIRD joint inventor, if any \_\_\_\_\_  
Citizenship \_\_\_\_\_ Residence \_\_\_\_\_

Post Office Address (If different) \_\_\_\_\_

Third Inventor's signature: \_\_\_\_\_ Date: \_\_\_\_\_

Full name of FOURTH joint inventor, if any \_\_\_\_\_  
Citizenship \_\_\_\_\_ Residence \_\_\_\_\_

Post Office Address (If different) \_\_\_\_\_

Fourth Inventor's signature: \_\_\_\_\_ Date: \_\_\_\_\_

### PART B: Prior Foreign Application(s)

<u>Serial No.</u>	<u>Country</u>	<u>Day/Month/Year Filed</u>	<u>Priority Claimed</u>
P10-15968	Japan	28/01/1998	<input checked="" type="checkbox"/> Yes <input type="checkbox"/> No

### PART C: Claim for Benefit of Filing Date of Earlier U.S. Application(s)

<u>Serial No.</u>	<u>Filing Date</u>	<u>Status:</u>
		<input type="checkbox"/> Patented <input type="checkbox"/> Pending
		<input type="checkbox"/> Patented <input type="checkbox"/> Pending <input type="checkbox"/> Abandoned
		<input type="checkbox"/> Patented <input type="checkbox"/> Pending <input type="checkbox"/> Abandoned